

FORM PCT/1390 (Modified)  
(REV 11-98)

U.S. DEPARTMENT OF COMMERCE PATENT AND TRADEMARK OFFICE

ATTORNEY'S DOCKET NUMBER

**TRANSMITTAL LETTER TO THE UNITED STATES  
DESIGNATED/ELECTED OFFICE (DO/EO/US)  
CONCERNING A FILING UNDER 35 U.S.C. 371**

**35615.180**

U.S. APPLICATION NO. (IF KNOWN, SEE 37 CFR

**09/554843**

INTERNATIONAL APPLICATION NO.  
**PCT/FR98/02486**

INTERNATIONAL FILING DATE  
**20 November, 1998 (20.11.98)**

PRIORITY DATE CLAIMED  
**21 November, 1997 (21.11.97)**

TITLE OF INVENTION

**PROCESS FOR ENCODING SPEECH AND TERMINALS FOR IMPLEMENTING THE PROCESS**

APPLICANT(S) FOR DO/EO/US  
**Pierre NICAISE**

Applicant herewith submits to the United States Designated/Elected Office (DO/EO/US) the following items and other information:

1. ☒ This is a **FIRST** submission of items concerning a filing under 35 U.S.C. 371.
2. ☐ This is a **SECOND** or **SUBSEQUENT** submission of items concerning a filing under 35 U.S.C. 371.
3. ☒ This is an express request to begin national examination procedures (35 U.S.C. 371(f)) at any time rather than delay examination until the expiration of the applicable time limit set in 35 U.S.C. 371(b) and PCT Articles 22 and 39(1).
4. ☒ A proper Demand for International Preliminary Examination was made by the 19th month from the earliest claimed priority date.
5. ☒ A copy of the International Application as filed (35 U.S.C. 371 (c) (2))
  - a. ☐ is transmitted herewith (required only if not transmitted by the International Bureau).
  - b. ☒ has been transmitted by the International Bureau.
  - c. ☐ is not required, as the application was filed in the United States Receiving Office (RO/US).
6. ☐ A translation of the International Application into English (35 U.S.C. 371(c)(2)).
7. ☒ A copy of the International Search Report (PCT/ISA/210).
8. ☐ Amendments to the claims of the International Application under PCT Article 19 (35 U.S.C. 371 (c)(3))
  - a. ☐ are transmitted herewith (required only if not transmitted by the International Bureau).
  - b. ☐ have been transmitted by the International Bureau.
  - c. ☐ have not been made; however, the time limit for making such amendments has NOT expired.
  - d. ☐ have not been made and will not be made.
9. ☒ A translation of the amendments to the claims under PCT Article 19 (35 U.S.C. 371(c)(3)).
10. ☒ An oath or declaration of the inventor(s) (35 U.S.C. 371 (c)(4)).
11. ☒ A copy of the International Preliminary Examination Report (PCT/IPEA/409).
12. ☐ A translation of the annexes to the International Preliminary Examination Report under PCT Article 36 (35 U.S.C. 371 (c)(5)).

**Items 13 to 20 below concern document(s) or information included:**

13. ☒ An Information Disclosure Statement under 37 CFR 1.97 and 1.98.
14. ☐ An assignment document for recording. A separate cover sheet in compliance with 37 CFR 3.28 and 3.31 is included.
15. ☒ A **FIRST** preliminary amendment.
16. ☐ A **SECOND** or **SUBSEQUENT** preliminary amendment.
17. ☐ A substitute specification.
18. ☐ A change of power of attorney and/or address letter.
19. ☒ Certificate of Mailing by Express Mail
20. ☐ Other items or information:

APPLICATION NO. (IF KNOWN, SEE 37 CFR

09/554843

INTERNATIONAL APPLICATION NO.

PCT/FR98/02486

ATTORNEY'S DOCKET NUMBER

35615.180

The following fees are submitted:

**BASIC NATIONAL FEE ( 37 CFR 1.492 (a) (1) - (5) ) :**

- ☐ Neither international preliminary examination fee (37 CFR 1.482) nor international search fee (37 CFR 1.445(a)(2)) paid to USPTO and International Search Report not prepared by the EPO or JPO ..... \$970.00
- ☒ International preliminary examination fee (37 CFR 1.482) not paid to USPTO but International Search Report prepared by the EPO or JPO ..... \$840.00
- ☐ International preliminary examination fee (37 CFR 1.482) not paid to USPTO but international search fee (37 CFR 1.445(a)(2)) paid to USPTO ..... \$690.00
- ☐ International preliminary examination fee paid to USPTO (37 CFR 1.482) but all claims did not satisfy provisions of PCT Article 33(1)-(4) ..... \$670.00
- ☐ International preliminary examination fee paid to USPTO (37 CFR 1.482) and all claims satisfied provisions of PCT Article 33(1)-(4) ..... \$96.00

**ENTER APPROPRIATE BASIC FEE AMOUNT =**

\$840.00

Surcharge of \$130.00 for furnishing the oath or declaration later than ☐ 20 ☐ 30 months from the earliest claimed priority date (37 CFR 1.492 (e)).

\$0.00

CLAIMS-	NUMBER FILED	NUMBER EXTRA	RATE
Total claims	10 - 20 =	0	x \$18.00
Independent claims	- 3 =	0	x \$78.00

\$0.00

\$0.00

Multiple Dependent Claims (check if applicable). ☐

\$0.00

**TOTAL OF ABOVE CALCULATIONS =**

\$840.00

Reduction of 1/2 for filing by small entity, if applicable. Verified Small Entity Statement must also be filed (Note 37 CFR 1.9, 1.27, 1.28) (check if applicable). ☐

\$0.00

**SUBTOTAL =**

\$840.00

Processing fee of \$130.00 for furnishing the English translation later than ☐ 20 ☐ 30 months from the earliest claimed priority date (37 CFR 1.492 (f)).

\$0.00

**TOTAL NATIONAL FEE =**

\$840.00

Fee for recording the enclosed assignment (37 CFR 1.21(h)). The assignment must be accompanied by an appropriate cover sheet (37 CFR 3.28, 3.31) (check if applicable). ☐

\$0.00

**TOTAL FEES ENCLOSED =**

\$840.00

Amount to be:

refunded

\$

charged

\$

☒ A check in the amount of \$840.00 to cover the above fees is enclosed.

☐ Please charge my Deposit Account No. in the amount of to cover the above fees.  
A duplicate copy of this sheet is enclosed.

☒ The Commissioner is hereby authorized to charge any fees which may be required, or credit any overpayment to Deposit Account No. 07-1855 A duplicate copy of this sheet is enclosed.

**NOTE: Where an appropriate time limit under 37 CFR 1.494 or 1.495 has not been met, a petition to revive (37 CFR 1.137(a) or (b)) must be filed and granted to restore the application to pending status.**

SEND ALL CORRESPONDENCE TO:

Joseph M. Manak, Esq.  
Graham & James LLP  
885 Third Avenue - 21st fl.  
New York, NY 10022

SIGNATURE

Joseph M. Manak

NAME

33,013

REGISTRATION NUMBER

May 19, 2000

DATE

09/554843

422 Rec'd PCT/PTO 19 MAY 2000

IN THE UNITED STATES PATENT AND TRADEMARK OFFICE

**IN RE APPLICATION OF:**  
Pierre Nicaise

**GROUP ART UNIT:** To Be Assigned

**SERIAL NO.:** To be Assigned

**EXAMINER:** To be Assigned

**FILED:** Of even date

**ATTORNEY DOCKET NO.:** 35615.180

**FOR:** PROCESS FOR ENCODING  
SPEECH AND TERMINALS FOR  
IMPLEMENTING THE PROCESS

BOX PCT  
Assistant Commissioner for Patents  
Washington, D.C. 20231

Sir:

**PRELIMINARY AMENDMENT**

Preliminary to the examination of the above identified application, please  
amend the application as follows:

**In the Specification:**

Insert, at line 25, after "restored.", the following: "The article, "Speaker normalization algorithms for very low-rate speech coding", of Roucos et al, ICASSP 84, Vol. 1, 19-21 March 1984, page 1.11 to 1.14, XP002073267, SAN DIEGO, CA, US, and the patent DE-A-3 416 238 disclose the progressive adaptation of a library of voice signals to those of a new speaker. The differences found between these signals serve only for the purposes of this adaptation. The speech signals are thus deformed upon restoration while the library has not reached a new state of equilibrium. WO-A-94 18 668 relates to the transmission of speech by linear prediction and does not relate to voice recognition. Patent Abstracts of Japan, Vol. 010, No. 240, JP 61 071 730A discloses recognition, apart from the speech, of a voice characteristic but without indicating either its nature, or, in particular, the

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manner of determining this characteristic. Faithful reproduction of the speech cannot be ensured."

**In the Claims:**

Please cancel claim 10. Please add the following claims:

11. A terminal of claim 8 formed by a facsimile machine comprising means to insert the code words into a facsimile message.
12. A terminal of claim 9 formed by a facsimile machine comprising means to insert the code words into a facsimile message.

**REMARKS**

The specification has been amended at page 1, line 25, to provide a description of art cited in the preliminary examination report dated June 7, 1999, a copy of which is submitted herewith. The amendment to the specification is for clarification purposes and adds no new matter.

Claims 1-9, 11 and 12 appear in the application for the Examiner's review and consideration. Claim 10 has been canceled and new claims 11-12 have been added in order to delete the multiple dependencies in claims presented in the PCT application. The new claims are thus totally supported and no new matter has been added thereby.


Applicant respectfully submits that claims 1-9 and 11-12 are all in condition for allowance, an early notice of which would be appreciated.

**AUTHORIZATION**

No fee is believed to be due with the filing of this Preliminary Amendment. Should any fees be due, however, authorization is hereby provided to charge the required amount to Graham & James LLP Deposit Account No. 07-1855.

Respectfully, submitted,

Date: May 19, 2000

  
\_\_\_\_\_  
Joseph M. Manak  
Registration No. 33,013

**GRAHAM & JAMES LLP**

885 Third Avenue  
New York, NY 10022

212-848-1087

2000-05-19 14:00:00

AR 34

New page 1a filed in response to the first written opinion

1a

The article .Speaker normalization algorithms for very low-rate speech coding. of Roucos et al, ICASSP 84, Vol. 1, 19-21 March 1984, page 1.11 to 1.14, XP002073267, SAN DIEGO, CA, US, and the patent DE-A-3 416 238 disclose the progressive adaptation of a library of voice signals to those of a new speaker. The differences found between these signals serve only for the purposes of this adaptation.

The speech signals are thus deformed upon restoration while the library has not reached a new state of equilibrium.

WO-A-94 18 668 relates to the transmission of speech by linear prediction and does not relate to voice recognition.

Patent Abstracts of Japan, Vol. 010, No. 240, JP 61 071 730A discloses recognition, apart from the speech, of a voice characteristic but without indicating either its nature or, in particular, the manner of determining this characteristic.

Faithful reproduction of the speech cannot be ensured.

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the differences are encoded by secondary code words and pairs of primary and secondary codes are substituted for the units.

Thus the primary code words will effectively and compactly encode the largest part of the acoustic energy input, while the secondary code words will improve the fidelity of restoration but without greatly increasing the volume of code data since they relate to only a limited energy and a low number of bits makes it possible to encode this marginal energy modulating the primary, standard energy corresponding to the primary code words.

The invention also relates to a terminal for encoding speech signals, comprising means for inputting a sequence of acoustic units and transmitting it to comparator means arranged to compare the acoustic units successively with library references and thus select therein in each instance a specific primary code word of one of the references, the terminal being characterised by the fact that the comparator means are arranged to determine a difference between the input acoustic unit considered and the reference corresponding to the code word selected and to transmit this difference to transcoding means provided to supply, in response, a secondary code word corresponding to memory means arranged to associate the respective primary and secondary code words.

Finally, the invention relates to a terminal for decoding signals, comprising means for receiving signals representing primary code words of references of acoustic units in a library, and decoding means arranged to select certain ones of the references in the library according to the primary code words received and to control a transducer for restoration of speech signals accordingly, the terminal being characterised in that the decoding means are arranged also to decode secondary, correction code words associated with the primary code words, and to correct the selected voice references accordingly.

Although the process of the invention makes it necessary, all in all, to have an encoding terminal and a corresponding decoding terminal, each of these can be sold separately and the applicant thus intends to claim each one.

In particular, it is advantageous to provide a facsimile machine comprising means for inserting the code words into a facsimile message.

The invention will be better understood with the aid of the following description of a preferred embodiment of the process of the invention, with reference to the attached drawing in which:

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- Figure 1 schematically illustrates a transmission terminal and a reception terminal for voice signals for implementation of the process of the invention,
- Figure 2 shows the amplitude  $A$  of a speech signal over time  $t$ ,
- Figure 3 shows the amplitude  $K$  of lines of the spectrum of the speech signal 2 over the frequency  $F$ , and
- Figure 4 is a flow diagram showing the steps of the process.

The transmission terminal referenced 15 in Figure 1, which is in this case in the form of a portable handset for a radio communication network, comprises a microphone 26 for inputting the speech signal of its user, supplying an analogue/digital converter 27 connected at the output to a central microprocessor unit 28 associated with two libraries 11 and 12 of sound sequences or standardised acoustic units such as phonemes. The central unit 28 which encodes the speech controls a transmitter 29, in this case a radio transmitter, of which the transmissions are received by a reception circuit 30 of a speech restoration terminal 35. Figure 2 illustrates the amplitude of an acoustic unit according to time  $t$  and Figure 3 illustrates the amplitude  $K$  of the lines of the spectrum corresponding to a given moment.

More precisely, the central unit 28 comprises a comparator 16 to compare the acoustic units received from the converter 27 with library acoustic units. As explained in more detail with regard to Figure 4, the comparator 16 has the function of selecting the library reference which is the most similar to the signal currently analysed and also has the function of specifying this difference, ie. of providing a deviation value for each of the criteria which serve in the selection. This difference is transcoded by a transcoding circuit 17, in order to condense its expression into the form of a secondary code word which is stored in a memory 18 under the control of the comparator 16. This comparator, which has previously stored the primary code word in the memory 18, addresses and controls the writing therein so that the two code words, primary and secondary, are physically associated, as they are from a logic point of view, ie. that, for example, chaining is defined between the two memory zones containing them.

The reception terminal 35 comprises a central unit 33 carrying out inverse decoding of the speech to supply a loudspeaker 34. Two memories forming libraries 31 and 32, in this case external to the central unit 33, are connected to this central unit. The reception terminal 35 is in this case a standard terminal for receiving written messages, known as a pager, also arranged to receive voice messages. For the sake of clarity various standard dialling input, dialling transmission and data display circuits have not been shown.



The central unit 33 comprises a circuit 36 for addressing the libraries 31 and 32, the personalised library and primary library respectively, on the basis of code words received from the reception circuit 30. In response, a buffer circuit 37 receives, from the primary library 32, spectra of primary acoustic units and transmits them to a circuit 38 for modulation or composition of these spectra. The circuit 38 modulates these spectra according to the secondary code word associated with the primary code word read from the primary library 32. The circuit 38 thus combines the information of the primary and secondary code words to restore the speech signal initially captured (26). This combination can, for example, be an addition or multiplication of frequency lines followed by an inverse Fourier transformation or it can also relate directly to signal amplitudes. In this example, each type of restored acoustic unit is stored in the personalised memory 31 in order to use this latter directly if an identical pair of code words, primary and secondary, is later received. In one variation, the memory 31 could contain only modulation values which it would supply to the circuit 38 after addressing by a secondary code word.

The encoding and decoding operations will now be explained in more detail with reference to Figure 4.

In order to encode the voice a speech signal 26 is captured by the microphone 26 during a step 1 and, in this case, it is converted into a digital signal in the converter 27 in a step 2. The speech signal is then compared, in the central unit 28, with a plurality of reference signals from the library 11 in a step 3. The comparison is carried out instantly, in practice in a cyclic manner at high speed with respect to the speed of evolution of the analysed speech signal. This signal can be considered as being a sequence of acoustic units specific to a given language, such as vowels, diphthongs or hiatus, of which a representation has initially been placed in the library 11 and associated with a code word referred to as the primary code word, which is peculiar to each. When the library 11 and the libraries 12 and 32 mentioned hereinunder are being formed, a number of voice inputs from a single speaker or from several are carried out in order to form an average voice reference. However, in order to improve the efficacy of future recognition, a number of references are preferably stored (11, 12) for each acoustic unit in order to form a range of recognition permitting differences between speakers to be accommodated.

Each acoustic unit (Figure 2) corresponds to a particular evolution of the amplitude A or energy of the speech signal and has a duration able to vary according to the talking speed of the person speaking.

The step 3 thus consists of comparing the evolution of the amplitudes of the reference signals to that of the captured signal. In order to overcome variations in the talking speed, it is possible, for example, to consider only the succession of significant modulations of amplitude (variation of energy exceeding a threshold) without associating a time value notion therewith.

In Figure 2 the vertical arrows, of which there are eight, represent the amplitude of the extremes and thus form a signature, assumed, in this case, to represent a specific acoustic unit. If, leaving the time domain of Figure 2, the frequency domain is now considered, the Fourier transform of the momentary amplitude  $A$  of the signal at any point on the curve of Figure 2 can be shown by the spectrum of frequency lines of Figure 3. In practice, it is considered that the voice energy is essentially limited to three frequency bands located respectively around 0.1 kHz and two bands between about 1 and 3 kHz as well as 5 and 7 kHz respectively.

For this reason, if the curve of the amplitudes of Figure 2 is traced over time  $t$ , the amplitude  $K$  of each line of Figure 3 will be modulated according to the evolution of the amplitude  $A$  of the speech signal.

Thus, if the succession of the spectra of Figure 3 is stored, it is possible to restore the succession of the amplitudes  $A$  of the original signal by an inverse Fourier transform.

In order to limit the number of spectra to be processed, it is possible to effect only cyclic sampling, sufficiently close not to lose information. It is also possible to be limited to a narrow set of spectra of the extremes of amplitude which are shown by the eight arrows in Figure 2. If it is desired to limit the number of spectra further, it is even possible to retain only a single spectrum representing the average of all the spectra over the considered period of time of the acoustic unit or the average of the spectra of the extremes.

As explained hereinunder in more detail, the average spectrum, or the spectra of the speech signal captured will be compared with one or more counterpart spectra of reference speech signals in the library in order, on the one hand, to select the reference speech signal (acoustic unit) which is the most similar to the captured signal and, on the other hand, to produce a signal representing the difference between the spectrum or spectra of this latter and the spectrum or spectra of the selected reference signal. The difference signal is formed into a code word, referred to as a secondary code word, and is associated with the primary code word of the selected reference signal (recognised acoustic unit) and thus constitutes an additional item of information for modulation or correction of the standardised analogue signal which will be restored from the primary code word considered.

The primary code words of the acoustic units, successively selected as the voice sequence progresses, are stored in a step 4 in order to form a message encoded according to the standard of the library 11.

Furthermore, in a step 5 some of the acoustic units captured and recognised are processed more by analysing their frequency spectrum in detail, in this case in the frequency domain by an inverse Fourier transform, as explained above, step 6.

In a step 7, the spectrum of lines  $j$  of the acoustic unit of identity  $i$  concerned or the spectra representing its evolution over time  $t$  are compared with the spectrum or spectra of the acoustic unit selected in the library 11 and which is or are contained in the associated library 12. In this way, for the or each spectrum a series of weighting coefficients  $C_{ijt}$  ( $i$  = identity of phoneme,  $j$  = frequency rank of the line,  $t$  = time rank) is established, each indicating the amplitude or relative energy of each line  $j$  with respect to its counterpart in the library 12. In other words, these coefficients also represent, albeit indirectly, the relative difference  $(1 - C_{ijt})$  between the recognised acoustic unit and the corresponding reference in the library). The lines in each of the three bands actually correspond to a row of mini bands of adjacent frequencies, in which the voice energy is detected. The analysis in the frequency domain, which is selected in this case, thus provides a more detailed item of information than in the case of an analysis in the time domain of Figure 2 where only the momentary amplitude  $A$  is available.

Thus, in the case of Figures 2 and 3, the series above comprises twelve coefficients representing the twelve lines shown, so that a table of eight such series represents the acoustic unit through the eight extremes shown. Apart from the reduction of the table to a single series, it is possible to make provision to retain only a single average weighting coefficient for each of the three bands. If each coefficient is encoded on just 4 bits, the error will not exceed about 3%, which is amply sufficient to restore a voice timbre, especially since the correction signal represents little energy with respect to the normed signal which it corrects, so that the error viewed as a total is low.

It is thus possible in this case to associate with the primary code word of the acoustic unit selected, of the order of about a hundred bits ( $12 \times 8$ ) if each extreme is retained, or only 12 bits ( $4 \times 3$ ) for the three bands. As the voice timbre is particularly provided by the high frequency of the third band, it is even possible to transmit only the secondary, correction code word relating thereto.

In a step 8, the signal representing the difference in the spectra is transformed into a secondary code word representing the table or the series mentioned above. When the captured speech sequence ends, the primary code words of step 4 and the secondary code words of step 8 are associated one to one (step 9) then transmitted on a transmission network such as, for example, the switched telephone network or, in this case, a radio messaging network (step 10).

The called terminal 35 receives the message in a step 21 and, in a step 22, a primary library file 32 similar to the file of spectra 12, is read by the circuit 36 in order to extract therefrom the primary standardised spectra according to the primary code words. In a step 23 the secondary code words serve to modulate (38) the amplitudes or energies of the standardised lines read in the primary library 32 in order, in this way, to constitute the personalised library 31 of acoustic units, ie. comprising, in particular, the timbre of the captured voice. The acoustic units of the personalised library 31 are represented in digital form in the time domain after previous transformation by an inverse Fourier transform.

In a step 24, the primary code words received are read successively in order to restore the captured speech signal via the loudspeaker 34 (step 25). For this purpose the primary code words read the personalised library 31 which thus corresponds to the library 11 but which has been personalised by the features in the spectrum of the captured voice.

As stated above, the formation of the library 31 is optional and aims to store a correction for each primary code word, which avoids having to repeat the sending of the secondary code word when the same primary code word is transmitted several times. If, on the other hand, a secondary code word is transmitted systematically, this code word can evolve to follow the possible evolutions in the timbre. In this case the restored voice is both personalised and the evolution of the timbre over time is also restored.

It should also be noted that the analysis and restoration can generally relate to the whole audible frequency band from about 15 Hz to 15 kHz, even if, in practice, it can be limited to 8 kHz. The frequencies of the band from 4 to 8 kHz, cut for standard transmission by the telephone network, are in this case analysed and restored since the corresponding information is transmitted in the form of a remote command from the library 31 which already contains the lines at these frequencies, which avoids any explicit transmission thereof.

It should also be noted that if the analysis can relate to only a limited number of sufficiently characteristic frequency bands in the library 11, 12, the various signals to be restored, in library

32, contain all the lines initially input, ie. each cover, for example, a single-piece band of 15 Hz to 8 kHz.

As explained at the beginning, the invention can apply in cases not involving transmission, in order, for example to store locally a message which to be restored later, ie. this would be a tape recording function.

In another embodiment, not illustrated, the primary and secondary code words are associated with facsimile data to form a voice-data message. The message is input via the telephone usually associated with facsimile machines and is restored by the same means at the called facsimile machine. The code words emitted by a circuit such as 28 are inserted in a specific field of the message by a microprocessor managing the facsimile transmission protocol and in the same way are retrieved upon reception to be processed as explained above. It is thus possible to carry out voice annotation of a facsimile message, annotation which is transmitted, for example, as a facsimile header.

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## CLAIMS

- 1 Process for encoding speech formed from a sequence of acoustic units, in which the units are compared with library references associated with primary code words, the differences between the units and the references are determined, the differences are encoded by secondary code words and pairs of primary and secondary codes are substituted for the units.
- 2 Process according to claim 1, wherein since the comparison relates to the energies of spectra of lines of frequencies, weighting coefficients normed with respect to the energy of the reference lines are determined for the lines, and the said coefficients are integrated into the secondary code word.
- 3 Process according to claim 1, wherein the said difference is determined from a succession of spectra corresponding to a succession of amplitudes of the acoustic unit considered.
- 4 Process according to claim 3, wherein only the amplitudes corresponding to extremes are considered.
- 5 Process according to claim 2, wherein the said difference is determined from a single average spectrum of the acoustic unit considered.
- 6 Process according to claim 2, wherein the frequency comparison is limited to three frequency bands.
- 7 Process according to claim 6, wherein the weighting coefficient of the lines of each band is expressed by a single coefficient.
- 8 Terminal for encoding speech signals, comprising means (26, 27) for inputting a sequence of acoustic units and transmitting it to comparator means (16) arranged to compare successively the acoustic units with references in libraries (11, 12) and thus select therein in each instance a specific primary code word of one of the references, the terminal being characterised by the fact that the comparator means (16) are arranged to determine a difference between the input acoustic unit considered and the reference corresponding to the code word selected and to transmit this difference to transcoding means (17) provided to

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NEW PAGE 12 FILED IN RESPONSE TO THE FIRST WRITTEN OPINION

acoustic units with references in libraries (11, 12) and thus select therein in each instance a specific primary code word of one of the references, the terminal being characterised by the fact that the comparator means (16) are arranged to determine a difference between the input acoustic unit considered and the reference corresponding to the code word selected and to transmit this difference to transcoding means (17) provided to supply, in response, a secondary code word corresponding to memory means (18) arranged to associate the respective primary and secondary code words.

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- 9 Terminal for decoding speech signals, comprising means (30) for receiving signals representing primary code words of references of acoustic units in a library (32), and decoding means (33) arranged to select certain ones of the references in the library (32) according to the primary code words received and to control a transducer (34) for restoration of speech signals accordingly, the terminal being characterised in that the decoding means (33) are arranged also to decode secondary, correction code words associated with the primary code words, and to correct (38) the selected voice references accordingly.
  - 10 Terminal according to one of claims 8 and 9, formed by a facsimile machine comprising means to insert the code words into a facsimile message.

**ABSTRACT****Process for encoding speech and terminals  
for implementing the process.**

The process for encoding speech formed from a sequence of acoustic units consists of comparing the units with library references associated with primary code words and determining the differences between the units and the references, the differences are encoded by secondary code words to substitute pairs of primary and secondary codes for the units and, if the comparison relates to the energies of spectra of lines of frequencies, weighting coefficients normed with respect to the energy of the reference lines are determined for the lines, and the coefficients are integrated into the secondary code word.

Figure 1

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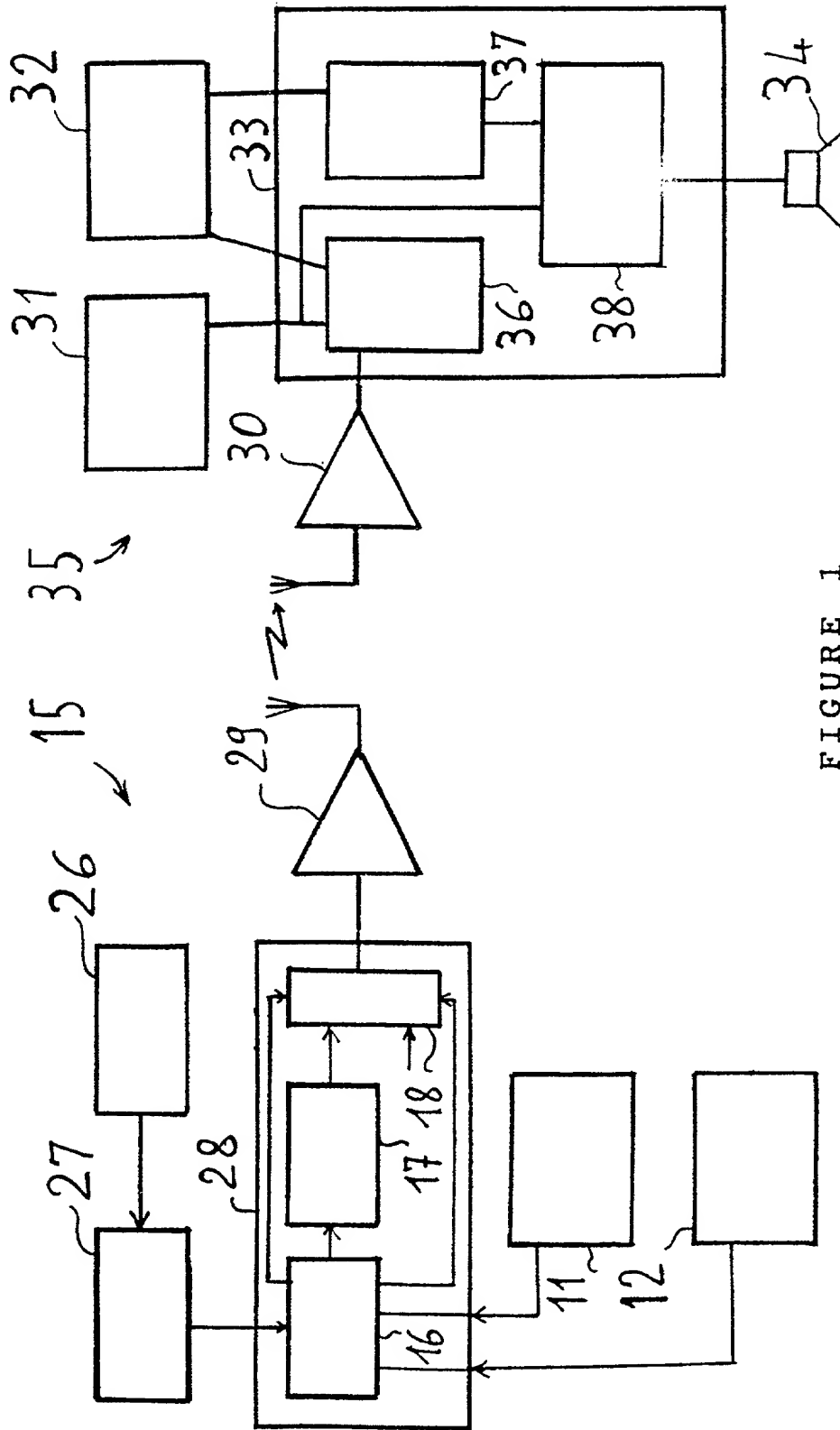


FIGURE 1

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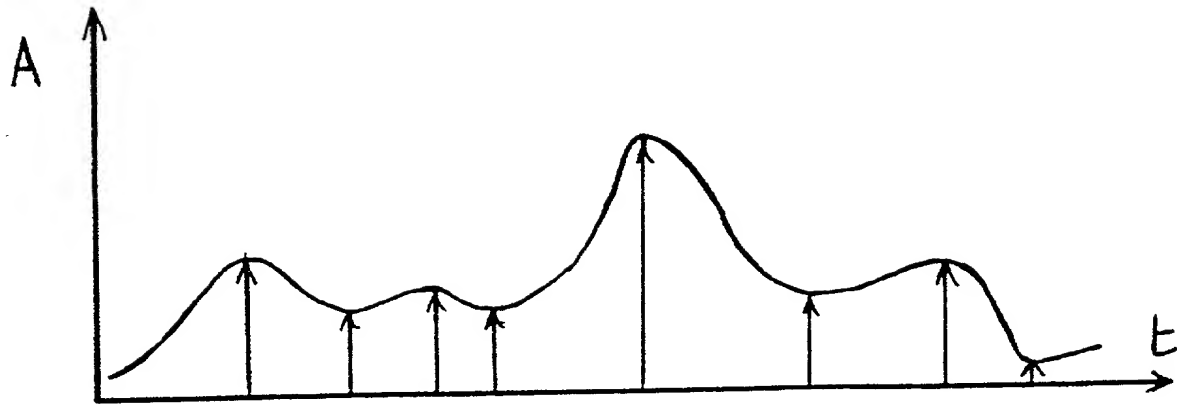
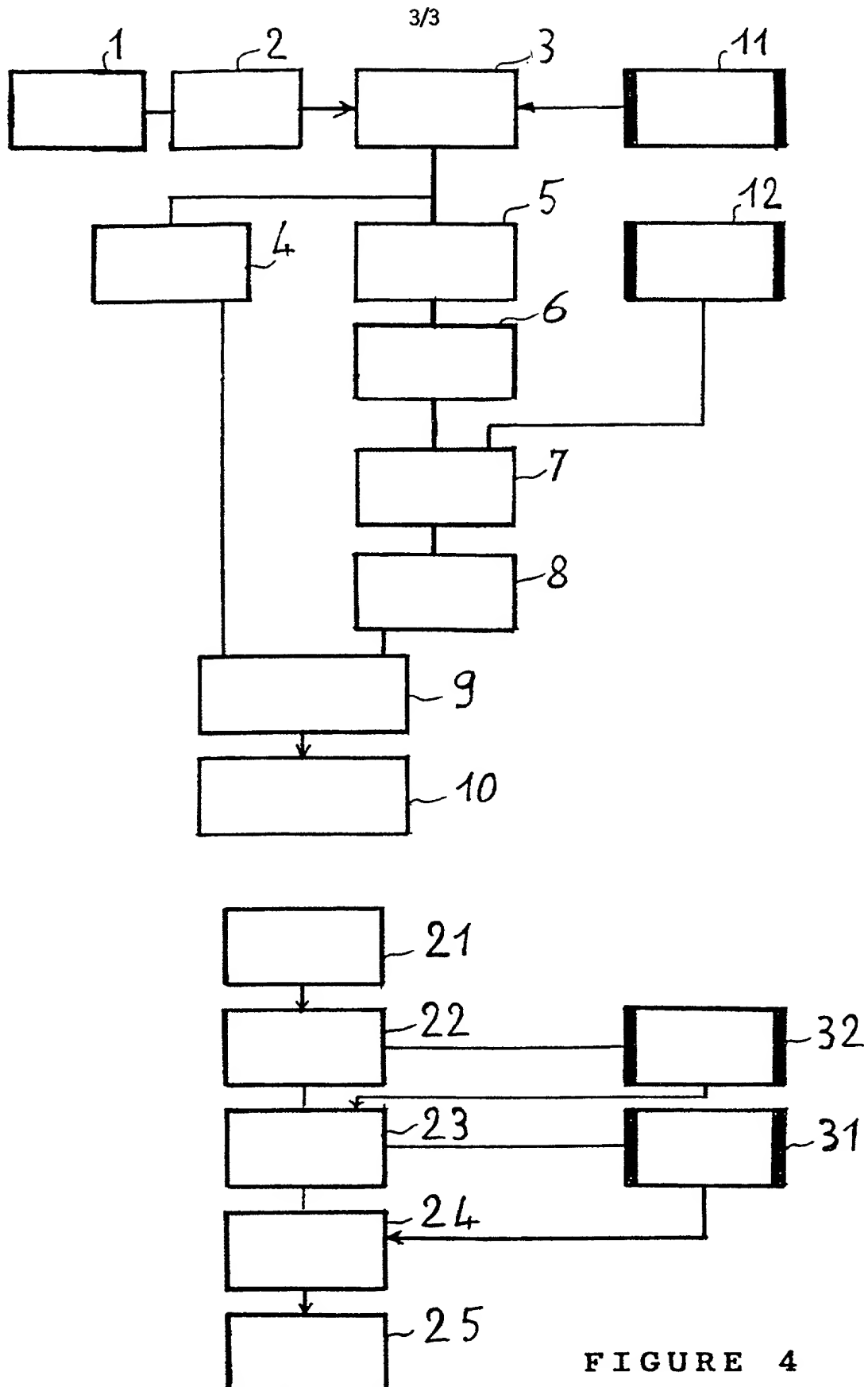


FIGURE 2



FIGURE 3



## DECLARATION FOR PATENT APPLICATION

As a below named inventor, I hereby declare that:

My residence, post office address and citizenship are as stated below next to my name.

I believe I am the original, first and sole inventor (if only one name is listed below) or an original, first and joint inventor (if plural names are listed below) of the subject matter which is claimed and for which a patent is sought on the invention entitled PROCESS FOR ENCODING SPEECH AND TERMINALS FOR

### IMPLEMENTING THE PROCESS

the specification of which is attached hereto unless the following box is checked:

☐ was filed on \_\_\_\_\_ as United States Application Serial Number \_\_\_\_\_ or PCT International Application Number \_\_\_\_\_ and was amended on \_\_\_\_\_ (if applicable).

I hereby state that I have reviewed and understand the contents of the above identified specification, including the claims, as amended by any amendment referred to above.

I acknowledge the duty to disclose information which is material to patentability as defined in 37 CFR § 1.56.

I hereby claim foreign priority benefits under 35 U.S.C. § 119(a)-(d) or § 365(b) of any foreign application(s) for patent or inventor's certificate, or § 365(a) of any PCT International application which designated at least one country other than the United States, listed below and have also identified below any foreign application for patent or inventor's certificate, or PCT International application having a filing date before that of the application on which priority is claimed.

#### Prior Foreign Application(s)

#### Priority Claimed

Yes No

<u>97-14667</u> (Number)	<u>FRANCE</u> (Country) (PCT)	<u>21 NOVEMBER 1997</u> (Day/Month/Year Filed)	<input checked="" type="checkbox"/>	<input type="checkbox"/>
<u>PCT/FR98/02486</u> (Number)	<u>PCT</u> (Country) (PCT)	<u>20 NOVEMBER 1998</u> (Day/Month/Year Filed)	<input checked="" type="checkbox"/>	<input type="checkbox"/>
_____ (Number)	_____ (Country) (PCT)	_____ (Day/Month/Year Filed)	<input type="checkbox"/>	<input type="checkbox"/>

I hereby claim the benefit under 35 U.S.C. § 119(e) of any United States provisional application(s) listed below.

\_\_\_\_\_  
(Application Number) (Filing Date)

\_\_\_\_\_  
(Application Number) (Filing Date)

I hereby claim the benefit under 35 U.S.C. § 120 of any United States application(s), or § 365(c) of any PCT International application designating the United States, listed below and, insofar as the subject matter of each of the claims of this application is not disclosed in the prior United States or PCT International application in the manner provided by the first paragraph of 35 U.S.C. § 112, I acknowledge the duty to disclose information which is material to patentability as defined in 37 CFR § 1.56 which became available between the filing date of the prior application and the national or PCT International filing date of this application.

\_\_\_\_\_  
(Application Number) (Filing Date) (Status -- patented, pending, abandoned)

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(Application Number) (Filing Date) (Status -- patented, pending, abandoned)

I hereby appoint the following attorney(s) and/or agent(s) to prosecute this application and to transact all business in the Patent and Trademark Office connected therewith:

Albert L. Jacobs, Jr. Reg. No. 22,211  
Jesse D. Reingold Reg. No. 20,461  
Gerard F. Diebner Reg. No. 31,345  
Israel Nissenbaum Reg. No. 27,582

Michael J. Brown Reg. No. 37,100  
Adam B. Landa Reg. No. 35,236  
Joseph R. Keating Reg. No. 37,368  
Philip M Weiss Reg. No. 34,751

Address all correspondence to: Intellectual Property Group  
Graham & James LLP  
885 Third Avenue  
New York, New York 10022  
(212) 848-1000

I hereby declare that all statements made herein of my own knowledge are true and that all statements made on information and belief are believed to be true; and further that these statements were made with the knowledge that willful false statements and the like so made are punishable by fine or imprisonment, or both, under Section 1001 of Title 18 of the United States Code and that such willful false statements may jeopardize the validity of the application or any patent issued thereon.

Full name of sole or first inventor (given name, family name) Pierre NICAISE

Inventor's signature [Signature] Date May 2, 2000

Residence COYE LA FORET Citizenship French

Post Office Address 16 rue des Hêtres, 60580 Coyo la Forêt France

Full name of second joint inventor, if any (given name, family name) \_\_\_\_\_

Second Inventor's signature \_\_\_\_\_ Date \_\_\_\_\_

Residence \_\_\_\_\_ Citizenship \_\_\_\_\_

Post Office Address \_\_\_\_\_

Full name of third joint inventor, if any (given name, family name) \_\_\_\_\_

Third Inventor's signature \_\_\_\_\_ Date \_\_\_\_\_

Residence \_\_\_\_\_ Citizenship \_\_\_\_\_

Post Office Address \_\_\_\_\_

☐ Additional inventors are being named on separately numbered sheets attached hereto.